

Mumbai University

Question Paper

**[CBSGS – 60:40 PATTERN]
(APRIL – 2014)**

PAPER - II

DIGITAL

SIGNALS AND SYSTEMS

Time: 2 ½ Hours

Total Marks: 60

N.B.: (1) All Question are Compulsory.

(2) Make Suitable Assumptions Wherever Necessary And State The Assumptions Made.

(3) Answer To The Same Question Must Be Written Together.

(4) Number To The Right Indicates Marks.

(5) Draw Neat Labeled Diagrams Wherever Necessary.

(6) Use of Non – Programmable Calculator is allowed.

Q.1 ATTEMPT ANY TWO QUESTIONS: (10 MARKS)

(A) Explain in detail the various types of systems. (5)

(B) Check whether the given is Power Signal or Energy Signal and find its value (5)

$$x[n] = 3(-1)^n, n \geq 0 \\ = 0, n < 0$$

(C) Explain in detail with suitable examples the various properties of Fourier Transform. (5)

(D) Find the Fourier Transform of the Time Function (5)

$$f(t) = 5[u(t+3) + u(t+2) - u(t-2) - u(t-3)]$$

Q.2 ATTEMPT ANY TWO QUESTIONS: (10 MARKS)

(A) Find Inverse Laplace Transform of (5)

$$(i) F_1(S) = \frac{S^2+5}{S^3+2S^2+4S}$$

$$(ii) F_2(S) = \frac{3e^{-\frac{S}{3}}}{S^2(S^2+2)}$$

(B) A sinusoidal voltage $25\sin t$ is applied at the instant $t = 0$ to a series RL Circuit with $R = 5\Omega$ and $L = 1 H$. Determine $i(t)$ by using Laplace Transform method. (5)(C) The unit step of a network is $(1 - e^{-n})$. Determine the Impulse Response $h(t)$ of the network. (5)

(D) Find the Laplace Transform of (5)

$$(i) e^{-t}$$

$$(ii) e^{10t}$$

$$(iii) 2 - 2e^t + 0.5 \sin 4t$$

$$(iv) e^{-t} \sin 4t$$

$$(v) e^{2t} + 2te^{-2t} - t^2$$

Q.3 ATTEMPT ANY TWO QUESTIONS: (10 MARKS)

(A) Explain the following properties of z-transform: (5)

(i) Time-reversal

(ii) Time Shifting

(iii) Time Scaling

(iv) Differentiation

(v) Convolution

(B) Determine the causal sequence $x(n)$ for $X(z)$ given by $X(z) = \frac{1+2z^{-1}}{1-2z^{-1}+4z^{-2}}$ (5)(C) Determine the causal signal having z-transform $X(z) = \frac{z^2+z}{(z-\frac{1}{2})^3(z-\frac{1}{4})}$ for the region of convergence (5)

$$|Z| > \frac{1}{2}$$

(D) For a low pass RC network, $R = 1 M\Omega$ and $C = 1\mu F$. Determine the output response for n in the range $C \leq n \leq 3$ when input has a step response of magnitude 2 V and the sampling frequency $f_1 = 50 Hz$. (5)

[TURN OVER]

Q.4 ATTEMPT ANY TWO QUESTIONS: (10 MARKS)

- (A) Explain the following properties of a Digital Signal Processing System: (5)
- (i) Linearity
 - (ii) Time-Invariance
 - (iii) Causality
 - (iv) Stability
 - (v) Bounded Input Bounded Output Stability
- (B) Consider a causal and stable LTI system whose input $x(n]$ and output $y(n]$ are related through the second order difference equation $y(n) - \frac{1}{12}y(n-1) - \frac{1}{12}y(n-2) = x(n]$ (5)
- (C) Determine the impulse response for the systems given by the following difference equations: (5)
- (i) $y(n) + 3y(n-1) + 2y(n-2) = 2x(n) - x(n-1)$
 - (ii) $y(n) = x(n) + 3x(n-1) - 4x(n-2) + 2x(n-3)$
- (D) Compute the response of the system $y(n) = 0.7y(n-1) - 0.12y(n-2) + x(n-1) + x(n-2)$ (5) to the input $x(n) = ny(n]$.

Q.5 ATTEMPT ANY TWO QUESTIONS: (10 MARKS)

- (A) Find the Circular Periodic Convolution using DFT and IDFT of the two sequences: (5)
- $x(n) = \{1, 1, 2, 2\}$ and $h(n) = \{1, 2, 3, 4\}$
- (B) Compute the Circular Periodic Convolution Graphically of the two sequences: (5)
- $x(n) = \delta(n) + \delta(n-1) - \delta(n-2) - \delta(n-3)$ and $h(n) = \delta(n) - \delta(n-2) + \delta(n-4)$
- (C) Given $x(n) = \{1, 2, 3, 4, 5, 6, 7, 8\}$. Find DFT and IDFT using FFT Algorithm. (5)
- (D) An FIR Digital Filter has the unit Impulse Response Sequence, $h(n) = \{2, 2, 1\}$. Determine the output sequence in response to the Input Sequence $x(n) = \{3, 0, -2, 0, 2, 1, 0, -2, -1, 0\}$ using the overlap-add convolution Method. (5)

Q.6 ATTEMPT ANY TWO QUESTIONS: (10 MARKS)

- (A) A low pass filter has the desired response as given below (5)
- $$H_e(e^{j\omega}) = e^{-j3\omega} \quad 0 \leq \omega \leq \frac{\pi}{2}$$
- $$= 0 \quad \frac{\pi}{2} \leq \omega \leq \pi$$
- Determine the filter coefficient $h(n)$ for $M = 7$, using Type-I frequency sampling technique.
- (B) Determine the unit sample response of the Ideal Low Pass Filter? Why is it not realizable? (5)
- (C) Design a High-Pass Digital FIR filter using Kaiser windows satisfying the specification given below. (5)
- Passband cut-off frequency, $f_p = 3200\text{Hz}$, stopband cut-off frequency, $f_s = 1600\text{Hz}$, passband ripple, $A_p = 0.1\text{dB}$, stopband attenuation, $A_s = 40\text{dB}$ and sampling frequency, $F = 10000\text{Hz}$.
- (D) An analog filter has the following system function. Convert this filter into a digital filter using backward difference for the derivative. $H(S) = \frac{1}{(S+0.1)^2 + 9}$ (5)